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- (71) Applicant: **MOTOROLA INC.** [US/US]: 1303 East Algonquin Road, Schaumburg, IL 60196 (US).
- (72) Inventors: **XU, Shuzham, J.**: Apartment 39N, 39 Reading Road, Edison, NJ 08817 (US). **TEICHER, Haim**; 19 Lincolnshire Drive, Manalapan, NJ 07726 (US).
- (74) Agents: **MANCINI, Brian et al.**: AN475, 600 North US Highway 45, Libertyville, IL 60048 (US).
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(54) Title: ITERATION TERMINATING USING QUALITY INDEX CRITERIA OF TURBO CODES

(57) Abstract: A decoder dynamically terminates iteration calculations in the decoding of a received convolutionally coded signal using quality index criteria. In a turbo decoder with two recursion processors connected in an iterative loop, at least one additional recursion processor is coupled in parallel at the inputs of at least one of the recursion processors. All of the recursion processors perform concurrent iterative calculations on the signal. The at least one additional recursion processor calculates a quality index of the signal for each iteration. A controller terminates the iterations when the measure of the quality index criteria exceeds a predetermined level.

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## ITERATION TERMINATING USING QUALITY INDEX CRITERIA OF TURBO CODES

### FIELD OF THE INVENTION

This invention relates generally to communication systems, and more  
5 particularly to a decoder for use in a receiver of a convolutionally coded  
communication system.

### BACKGROUND OF THE INVENTION

Convolutional codes are often used in digital communication systems to  
protect transmitted information from error. Such communication systems include  
10 the Direct Sequence Code Division Multiple Access (DS-CDMA) standard IS-95  
and the Global System for Mobile Communications (GSM). Typically in these  
systems, a signal is convolutionally coded into an outgoing code vector that is  
transmitted. At a receiver, a practical soft-decision decoder, such as a Viterbi  
decoder as is known in the art, uses a trellis structure to perform an optimum  
15 search for the maximum likelihood transmitted code vector.

More recently, turbo codes have been developed that outperform  
conventional coding techniques. Turbo codes are generally composed of two or  
more convolutional codes and turbo interleavers. Turbo decoding is iterative and  
uses a soft output decoder to decode the individual convolutional codes. The  
20 soft output decoder provides information on each bit position which helps the soft  
output decoder decode the other convolutional codes. The soft output decoder is  
usually a MAP (maximum *a posteriori*) or soft output Viterbi algorithm (SOVA)  
decoder.

Turbo coding is efficiently utilized to correct errors in the case of  
25 communicating over an added white Gaussian noise (AWGN) channel.  
Intuitively, there are a few ways to examine and evaluate the error correcting  
performance of the turbo decoder. One observation is that the magnitude of log-

likelihood ratio (LLR) for each information bit in the iterative portion of the decoder increases as iterations go on. This improves the probability of the correct decisions. The LLR magnitude increase is directly related to the number of iterations in the turbo decoding process. However, it is desirable to reduce the  
5 number of iterations to save calculation time and circuit power. The appropriate number of iterations (stopping criteria) for a reliably turbo decoded block varies as the quality of the incoming signal and the resulting number of errors incurred therein. In other words, the number of iterations needed is related to channel conditions, where a more noisy environment will need more iterations to correctly  
10 resolve the information bits and reduce error.

One prior art stopping criteria utilizes a parity check as an indicator to stop the decoding process. A parity check is straightforward as far as implementation is concerned. However, a parity check is not reliable if there are a large number of bit errors. Another type of criteria for the turbo decoding iteration stop is the  
15 LLR (log-likelihood-ratio) value as calculated for each decoded bit. Since turbo decoding converges after a number of iterations, the LLR of a data bit is the most direct indicator index for this convergence. One way this stopping criteria is applied is to compare LLR magnitude to a certain threshold. However, it can be difficult to determine the proper threshold as channel conditions are variable.  
20 Still other prior art stopping criteria measure the entropy or difference of two probability distributions, but this requires much calculation.

There is a need for a decoder that can determine the appropriate stopping point for the number of iterations of the decoder in a reliable manner. It would also be of benefit to provide the stopping criteria without a significant increase in  
25 calculation complexity.

## BRIEF DESCRIPTION OF THE DRAWINGS

FIG. 1 shows a trellis diagram used in soft output decoder techniques as are known in the prior art;

FIG. 2 shows a simplified block diagram for turbo encoding as is known in the prior art;

FIG. 3 shows a simplified block diagram for a turbo decoder as is known in the prior art;

5        FIG. 4 shows a simplified block diagram for a turbo decoder with an iterative quality index criteria, in accordance with the present invention;

FIG. 5 shows simplified block diagram for the Viterbi decoder as used in FIG. 4;

10       FIG. 6 shows a graphical representation of the improvement provided by the hard quality index of the present invention;

FIG. 7 shows a graphical representation of the improvement provided by the soft quality index of the present invention;

FIG. 8 shows another graphical representation of the improvement provided by the present invention;

15       FIG. 9 shows a graphical representation of the improvement provided by the hard intrinsic SNR index of the present invention;

FIG. 10 shows a graphical representation of the improvement provided by the soft intrinsic SNR index of the present invention; and

20       FIG. 11 shows a method for turbo decoding, in accordance with the present invention.

#### DETAILED DESCRIPTION OF THE PREFERRED EMBODIMENT

The present invention provides a turbo decoder that dynamically utilizes the virtual (intrinsic) SNR as a quality index stopping criteria of the in-loop data stream at the input of each constituent decoder stage, as the loop decoding iterations proceed. This quality index is used as a stopping criteria to determine the number of iterations needed in the decoder. Advantageously, by limiting the number of calculations to be performed in order to decode bits reliably, the present invention conserves power in the communication device and saves calculation complexity.

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Typically, block codes, convolutional codes, turbo codes, and others are graphically represented as a trellis as shown in FIG. 1, wherein a four state, five section trellis is shown. For convenience, we will reference  $M$  states per trellis section (typically  $M$  equals eight states) and  $N$  trellis sections per block or frame (typically  $N=5000$ ). Maximum *a posteriori* type decoders (log-MAP, MAP, max-log-MAP, constant-log-MAP, etc.) utilize forward and backward generalized Viterbi recursions or soft output Viterbi algorithms (SOVA) on the trellis in order to provide soft outputs at each section, as is known in the art. The MAP decoder minimizes the decoded bit error probability for each information bit based on all received bits.

Because of the Markov nature of the encoded sequence (wherein previous states cannot affect future states or future output branches), the MAP bit probability can be broken into the past (beginning of trellis to the present state), the present state (branch metric for the current value), and the future (end of trellis to current value). More specifically, the MAP decoder performs forward and backward recursions up to a present state wherein the past and future probabilities are used along with the present branch metric to generate an output decision. The principles of providing hard and soft output decisions are known in the art, and several variations of the above described decoding methods exist. Most of the soft input-soft output (SISO) decoders considered for turbo codes are based on the prior art optimal MAP algorithm in a paper by L.R. Bahl, J. Cocke, F. Jelinek, and J. Raviv entitled "Optimal Decoding of Linear Codes for Minimizing Symbol Error Rate", IEEE Transactions on Information Theory, Vol. IT-20, March 1974, pp. 284-7 (BCJR algorithm).

FIG. 2 shows a typical turbo coder that is constructed with interleavers and constituent codes which are usually systematic convolutional codes, but can be block codes also. In general, a turbo encoder is a parallel concatenation of two recursive systemic convolutional encoders (RSC) with an interleaver (int) between them. The output of the turbo encoding is generated by multiplexing (concatenating) the information bits  $m_i$  and the parity bits  $p_i$  from the two encoders, RSC1 and RSC2. Optionally, the parity bits can be punctured as is

known in the art to increase code rate (i.e., a throughput of 1/2). The turbo encoded signal is then transmitted over a channel. Noise,  $n_i$ , due to the AWGN nature of the channel becomes added to the signal,  $x_i$ , during transmission. The noise variance of the AWGN can be expressed as  $\sigma^2 = N_0/2$ , where  $N_0/2$  is the two sided noise power spectrum density. The noise increases the likelihood of bit errors when a receiver attempts to decode the input signal,  $y_i (= x_i + n_i)$ , to obtain the original information bits  $m_i$ . Correspondingly, noise affects the transmitted parity bits to provide a signal  $t_i = p_i + n_i$ .

FIG. 3 shows a typical turbo decoder that is constructed with interleavers, de-interleavers, and decoders. The mechanism of the turbo decoder regarding extrinsic information  $L_{e1}$ ,  $L_{e2}$ , interleaver (int), de-interleaver (deint), and the iteration process between the soft-input, soft-output decoder sections SISO1 and SISO2 follow the Bahl algorithm. Assuming zero decoder delay in the turbo decoder, the first decoder (SISO1) computes a soft output from the input signal bits,  $y_i$ , and the *a priori* information ( $L_a$ ), which will be described below. The soft output is denoted as  $L_{e1}$ , for extrinsic data from the first decoder. The second decoder (SISO2) is input with interleaved versions of  $L_{e1}$  (the *a priori* information from  $L_a$ ), the input signal bits  $y_i$ . The second decoder generates extrinsic data,  $L_{e2}$ , which is deinterleaved to produce  $L_a$  which is fed back to the first decoder, and a soft output (typically a MAP LLR) provide a soft output of the original information bits  $m_i$ . Typically, the above iterations are repeated for a fixed number of times (usually sixteen) for each bit until all the input bits are decoded.

MAP algorithms minimize the probability of error for an information bit given the received sequence, and they also provide the probability that the information bit is either a 1 or 0 given the received sequence. The prior art BCJR algorithm provides a soft output decision for each bit position (trellis section of FIG. 1) wherein the influence of the soft inputs within the block is broken into contributions from the past (earlier soft inputs), the present soft input, and the future (later soft inputs). The BCJR decoder algorithm uses a forward and a backward generalized Viterbi recursion on the trellis to arrive at an optimal soft output for each trellis section (stage). These *a posteriori* probabilities, or more

commonly the log-likelihood ratio (LLR) of the probabilities, are passed between SISO decoding steps in iterative turbo decoding. The LLR for each information bit is

$$\text{LLR}(b_k) = \frac{\sum_{n \in \mathcal{N}_1} \alpha_k(n) \gamma_k(n, m) \beta_k(m)}{\sum_{n \in \mathcal{N}_0} \alpha_k(n) \gamma_k(n, m) \beta_k(m)} \quad (1)$$

- 5 for all bits in the decoded sequence ( $k = 1$  to  $M$ ). In equation (1), the probability that the decoded bit is equal to 1 (or 0) in the trellis given the received sequence is composed of a product of terms due to the Markov property of the code. The Markov property states that the past and the future are independent given the present. The present,  $\gamma_k(n, m)$ , is the probability of being in state  $m$  at time  $k$  and generating the symbol  $\gamma_k$  when the previous state at time  $k-1$  was  $n$ . The present  
10 plays the function of a branch metric. The past,  $\alpha_k(m)$ , is the probability of being in state  $m$  at time  $k$  with the received sequence  $\{y_1, \dots, y_k\}$ , and the future,  $\beta_k(m)$ , is the probability of generating the received sequence  $\{y_{k+1}, \dots, y_N\}$  from state  $m$  at time  $k$ . The probability  $\alpha_k(m)$  can be expressed as function of  $\alpha_{k-1}(m)$  and  
15  $\gamma_k(n, m)$  and is called the forward recursion

$$\alpha_k(m) = \sum_{n=0}^{M-1} \alpha_{k-1}(n) \gamma_k(n, m) \quad m = 0, \dots, M-1, \quad (2)$$

where  $M$  is the number of states. The reverse or backward recursion for computing the probability  $\beta_k(n)$  from  $\beta_{k+1}(n)$  and  $\gamma_k(n, m)$  is

$$\beta_k(n) = \sum_{m=0}^{M-1} \beta_{k+1}(m) \gamma_k(n, m) \quad n = 0, \dots, M-1. \quad (3)$$

- 20 The overall *a posteriori* probabilities in equation (2) are computed by summing over the branches in the trellis  $B^1$  ( $B^0$ ) that correspond to the information bit being 1 (or 0).

- The LLR in equation (1) requires both the forward and reverse recursions to be available at time  $k$ . In general, the BCJR method for meeting this  
25 requirement is to compute and store the entire reverse recursion using a fixed

number of iterations, and recursively compute  $\alpha_k(m)$  and  $La_k$  from  $k = 1$  to  $k = N$  using  $\alpha_{k-1}$  and  $\beta_k$ .

The performance of turbo decoding is affected by many factors. One of the key factors is the number of iterations. As a turbo decoder converges after a few iterations, more iterations after convergence will not increase performance significantly. Turbo codes will converge faster under good channel conditions requiring a fewer number of iterations to obtain good performance. The number of iterations performed is directly proportional to the number of calculations needed and it will affect power consumption. Since power consumption is of great concern in the mobile and portable radio communication devices, there is an even higher emphasis on finding reliable and good iteration stopping criteria. Motivated by these reasons, the present invention provides an adaptive scheme for stopping the iteration process.

In the present invention, the number of iterations is defined as the total number of SISO decoding stages used (i.e. two iterations in one cycle). Accordingly, the iteration number counts from 0 to  $2N-1$ . Each decoding stage can be either MAP or SOVA. The key factor in the decoding process is to combine the extrinsic information into a SISO block. The final hard decision on the information bits is made according to the value of the LLR after iterations are stopped. The final hard bit decision is based on the LLR polarity. If the LLR is positive, decide +1, otherwise decide -1 for the hard output.

In the present invention, the in-loop signal-to-noise ratio (intrinsic SNR) is used as the iteration stopping criterion in the turbo decoder. Since SNR improves when more bits are detected correctly per iteration, the present invention uses a detection quality indicator that observes the increase in signal energy relative to the noise as iterations go on.

FIG. 4 shows a turbo decoder with at least one additional Viterbi decoder to monitor the decoding process, in accordance with the present invention. Although one Viterbi decoder can be used, two decoders give the flexibility to stop iterations at any SISO decoder. The Viterbi decoders are used because it is easy to analyze the Viterbi decoder to get the quality index. The Viterbi decoder



is just used to do the mathematics in the present invention, i.e. to derive the quality indexes and intrinsic SNR values. No real Viterbi decoding is needed. It is well known that MAP or SOVA will not outperform the conventional Viterbi decoder significantly if no iteration is applied. Therefore, the quality index also applies towards the performance of MAP and SOVA decoders. The error due to the Viterbi approximation to SISO (MAP or SOVA) will not accumulate since there is no change in the turbo decoding process itself. Note that the turbo decoding process remains as it is. The at least one additional Viterbi decoder is attached for analysis to generate the quality index and no decoding is actually needed.

In a preferred embodiment, two Viterbi decoders are used. In practice, where two identical RSC encoder are used, thus requiring identical SISO decoders, only one Viterbi decoder is needed, although two of the same decoders can be used. Otherwise, the two Viterbi decoders are different and they are both required. Both decoders generate an iteration stopping signal, and they act independently such that either decoder can signal a stop to iterations. The Viterbi decoders are not utilized in the traditional sense in that they are only used to do the mathematics and derive the quality indexes and intrinsic SNR values. In addition, since iterations can be stopped mid-cycle at any SISO decoder, a soft output is generated for the transmitted bits from the LLR of the decoder where the iteration is stopped.

The present invention utilizes the extrinsic information available in the iterative loop in the Viterbi decoder. For an AWGN channel, we have the following path metrics with the extrinsic information input:

$$p[Y | X] = \prod_{i=0}^{L-1} p[y_i | x_i] p[t_i | p_i] p[m_i]$$

where  $m_i$  is the transmitted information bit,  $x_i = m_i$  is the systematic bit, and  $p_i$  is the parity bit. With  $m_i$  in polarity form ( $1 \rightarrow +1$  and  $0 \rightarrow -1$ ), we rewrite the extrinsic information as

$$p[m_i] = \frac{e^{z_i}}{1 + e^{z_i}} = \frac{e^{z_i/2}}{e^{-z_i/2} + e^{z_i/2}}, \quad \text{if } m_i = +1$$

$$p[m_i] = \frac{1}{1 + e^{z_i}} = \frac{e^{-z_i/2}}{e^{-z_i/2} + e^{z_i/2}}, \quad \text{if } m_i = -1$$

$p[m_i]$  is the a priori information about the transmitted bits,  $z_i = \log \frac{p[m_i = +1]}{p[m_i = -1]}$  is the extrinsic information, or in general,

$$p[m_i] = \frac{e^{m_i z_i / 2}}{e^{-z_i/2} + e^{z_i/2}}$$

5

The path metric is thus calculated as

$$\begin{aligned} p[Y | X] &= \prod_{i=0}^{L-1} p[y_i | x_i] p[t_i | p_i] p[m_i] \\ &= \left( \frac{1}{\sqrt{2\pi}\sigma} \right)^L e^{-\frac{1}{2\sigma^2} \sum_{i=0}^{L-1} [(x_i - y_i)^2 + (p_i - t_i)^2]} \left( \prod_{i=0}^{L-1} \frac{1}{e^{-z_i/2} + e^{z_i/2}} \right) e^{\frac{1}{2} \sum_{i=0}^{L-1} m_i z_i} \end{aligned}$$

- 10 Note that  $e^{\frac{1}{2} \sum_{i=0}^{L-1} m_i z_i}$  is the correction factor introduced by the extrinsic information. And from the Viterbi decoder point of view, this correcting factor improves the path metric and thus improves the decoding performance. This factor is the improvement brought forth by the extrinsic information. The present invention introduces this factor as the quality index and the iteration stopping criteria for
- 15 turbo codes.

In particular, the turbo decoding quality index  $Q(\text{iter}, \{m_i\}, L)$  is:

$$Q(\text{iter}, \{m_i\}, L) = \sum_{i=0}^{L-1} m_i z_i$$

- where  $\text{iter}$  is the iteration number,  $L$  denote number of bits in each decoding block,  $m_i$  is the transmitted information bit, and  $z_i$  is the extrinsic information
- 20 generated after each small decoding step. More generally,

$$Q(\text{iter}, \{m_i\}, \{w_i\}, L) = \sum_{i=0}^{L-1} w_i m_i z_i$$

where  $w_i$  is a weighting function to alter performance. In a preferred embodiment,  $w_i$  is a constant of 1.

This index remains positive since typically  $z_i$  and  $m_i$  have the same polarity. In practice, the incoming data bits  $\{m_i\}$  are unknown, and the following index is used instead:

$$Q_H(iter, \{m_i\}, L) = \sum_{i=0}^{L-1} \hat{d}_i z_i$$

where  $\hat{d}_i$  is the hard decision as extracted from the LLR information. That is  $\hat{d}_i = \text{sign}\{L_i\}$  with  $L_i$  denoting the LLR value. The following soft output version of the quality index can also be used for the same purpose:

$$Q_S(iter, \{m_i\}, L) = \sum_{i=0}^{L-1} L_i z_i$$

or more generally

$$Q_S(iter, \{m_i\}, \{w_i\}, L) = \sum_{i=0}^{L-1} w_i L_i z_i$$

Note that these indexes are extremely easy to generate and require very little hardware. In addition, these indexes have virtually the same asymptotic behavior and can be used as a good quality index for the turbo decoding performance evaluation and iteration stopping criterion.

The behavior of these indexes is that they increase very quickly for the first a few iterations and then they approach an asymptote of almost constant value. As can be seen from simulation results below, this asymptotic behavior describes the turbo decoding process well and serves as a quality monitor of the turbo decoding process. In operation, the iterations are stopped if this index value crosses the knee of the asymptote.

The iterative loop of the turbo decoder increases the magnitude of the LLR such that the decision error probability will be reduced. Another way to look at it is that the extrinsic information input to each decoder is virtually improving the SNR of the input sample streams. The following analysis is presented to show that what the extrinsic information does is to improve the virtual SNR to each

constituent decoder. This helps to explain how the turbo coding gain is reached. Analysis of the incoming samples is also provided with the assistance of the Viterbi decoder as described before.

The path metric equation of the attached additional Viterbi decoders is

$$p[Y | X] = \left(\frac{1}{\sqrt{2\pi}\sigma}\right)^L e^{-\frac{1}{2\sigma^2} \sum_{i=0}^{L-1} ((x_i - y_i)^2 + (p_i - t_i)^2)} \left(\prod_{i=0}^{L-1} \frac{1}{e^{-z_i/2} + e^{z_i/2}}\right) e^{\frac{1}{2} \sum_{i=0}^{L-1} m_i z_i}$$

Expansion of this equation gives

$$\begin{aligned} p[Y | X] &= \left(\frac{1}{\sqrt{2\pi}\sigma}\right)^{2L} \left(\prod_{i=0}^{L-1} \frac{1}{e^{-z_i/2} + e^{z_i/2}}\right) \cdot e^{-\frac{1}{2\sigma^2} \sum_{i=0}^{L-1} (x_i^2 + y_i^2)} e^{-\frac{1}{2\sigma^2} \sum_{i=0}^{L-1} (t_i^2 + p_i^2)} e^{\frac{1}{2\sigma^2} \sum_{i=0}^{L-1} (2x_i y_i + 2t_i p_i)} e^{\frac{1}{2} \sum_{i=0}^{L-1} x_i z_i} \\ &= \left(\frac{1}{\sqrt{2\pi}\sigma}\right)^{2L} \left(\prod_{i=0}^{L-1} \frac{1}{e^{-z_i/2} + e^{z_i/2}}\right) \cdot e^{-\frac{1}{2\sigma^2} \sum_{i=0}^{L-1} (x_i^2 + y_i^2)} e^{-\frac{1}{2\sigma^2} \sum_{i=0}^{L-1} (t_i^2 + p_i^2)} e^{\frac{1}{\sigma^2} \sum_{i=0}^{L-1} (x_i y_i + t_i p_i)} e^{\frac{1}{2} \sum_{i=0}^{L-1} x_i z_i} \end{aligned}$$

Looking at the correlation term, we get the following factor

$$\begin{aligned} &\frac{1}{\sigma^2} \sum_{i=0}^{L-1} (x_i y_i + \frac{\sigma^2}{2} x_i z_i) + \frac{1}{\sigma^2} \sum_{i=0}^{L-1} t_i p_i \\ &= \frac{1}{\sigma^2} \sum_{i=0}^{L-1} x_i (y_i + \frac{\sigma^2}{2} z_i) + \frac{1}{\sigma^2} \sum_{i=0}^{L-1} t_i p_i \\ &= \frac{1}{\sigma^2} \sum_{i=0}^{L-1} \{x_i (y_i + \frac{\sigma^2}{2} z_i) + t_i p_i\} \end{aligned}$$

For the Viterbi decoder, to search for the minimum Euclidean distance is the same process as searching for the following maximum correlation.

$$\frac{1}{\sigma^2} \sum_{i=0}^{L-1} \{x_i (y_i + \frac{\sigma^2}{2} z_i) + t_i p_i\}$$

or equivalently, the input data stream to the Viterbi decoder is  $\{(y_i + \frac{\sigma^2}{2} z_i), t_i\}$ ,

which is graphically depicted in FIG. 5.

Following the standard signal-to-noise ratio calculation formula

$$SNR = \frac{(E[y_i | x_i])^2}{\sigma^2}$$

and given the fact that  $y_i = x_i + n_i$  and  $t_i = p_i + n_i$  (where  $p_i$  are the parity bits of the incoming signal), we get SNR for the input data samples into the constituent  
5 decoder as

$$\begin{aligned} SNR(x_i, y_i, iter) &= \frac{(E[y_i + \frac{\sigma^2}{2} z_i | x_i])^2}{\sigma^2} \\ &= \frac{(E[x_i + n_i + \frac{\sigma^2}{2} z_i | x_i])^2}{\sigma^2} \\ &= \frac{(x_i + \frac{\sigma^2}{2} z_i)^2}{\sigma^2} \\ &= \frac{x_i^2}{\sigma^2} + x_i z_i + \frac{\sigma^2}{4} z_i^2 \end{aligned}$$

10

Notice that the last two terms are correction terms due to the extrinsic information input. The SNR for the input parity samples are

$$SNR(p_i, t_i, iter) = \frac{(E[t_i | p_i])^2}{\sigma^2} = \frac{(E[p_i + n_i | p_i])^2}{\sigma^2} = \frac{p_i^2}{\sigma^2}$$

15

Now it can be seen that the SNR for each received data samples are changing as iterations go on because the input extrinsic information will increase the virtual or intrinsic SNR. Moreover, the corresponding SNR for each parity sample will not be affected by the iteration. Clearly, if  $x_i$  has the same sign as  $z_i$ , we have

$$20 \quad SNR(x_i, y_i, iter) = \frac{(x_i + \frac{\sigma^2}{2} z_i)^2}{\sigma^2} \geq \frac{x_i^2}{\sigma^2} = SNR(x_i, y_i, iter = 0)$$

This shows that the extrinsic information increased the virtual SNR of the data stream input to each constituent decoder.

The average SNR for the whole block is

$$\begin{aligned}
 \text{AverageSNR}(iter) &= \frac{1}{2L} \left\{ \sum_{i=0}^{L-1} \text{SNR}(x_i, y_i, iter) + \sum_{i=0}^{L-1} \text{SNR}(p_i, t_i, iter) \right\} \\
 &= \frac{1}{2L} \left\{ \sum_{i=0}^{L-1} \frac{x_i^2}{\sigma^2} + \sum_{i=0}^{L-1} \frac{p_i^2}{\sigma^2} \right\} + \frac{1}{2L} \left\{ \sum_{i=0}^{L-1} x_i z_i + \frac{\sigma^2}{4} \sum_{i=0}^{L-1} z_i^2 \right\} \\
 &= \text{AverageSNR}(0) + \frac{1}{2L} Q(iter, \{m_i\}, L) + \frac{\sigma^2}{4} \left( \frac{1}{2L} \sum_{i=0}^{L-1} z_i^2 \right)
 \end{aligned}$$

at each iteration stage.

If the extrinsic information has the same sign as the received data samples and if the magnitudes of the  $z_i$  samples are increasing, the average SNR of the whole block will increase as the number of iteration increases. Note that the second term is the original quality index, as described previously, divided by the block size. The third term is directly proportional to the average of magnitude squared of the extrinsic information and is always positive. This intrinsic SNR expression will have the similar asymptotic behavior as the previously described quality indexes and can also be used as decoding quality indicator. Similar to the quality indexes, more practical intrinsic SNR values are:

$$\text{AverageSNR}_H(iter) = \text{StartSNR} + \frac{1}{2L} Q_H(iter, \{m_i\}, L) + \frac{\sigma^2}{4} \left( \frac{1}{2L} \sum_{i=0}^{L-1} z_i^2 \right),$$

or a corresponding soft copy of it

$$\text{AverageSNR}_S(iter) = \text{StartSNR} + \frac{1}{2L} Q_S(iter, \{m_i\}, L) + \frac{\sigma^2}{4} \left( \frac{1}{2L} \sum_{i=0}^{L-1} z_i^2 \right)$$

where *StartSNR* denotes the initial SNR value that starts the decoding iterations. Optionally, a weighting function can be used here as well. Only the last two terms are needed to monitor the decoding quality. Note also that the normalization constant in the previous intrinsic SNR expressions has been ignored.

In review, the present invention provides a decoder that dynamically terminates iteration calculations in the decoding of a received convolutionally coded signal using quality index criteria. The decoder includes a standard turbo decoder with two recursion processors connected in an iterative loop. A novel aspect of the inventions is having at least one additional recursion processor coupled in parallel at the inputs of at least one of the recursion processors. Preferably, the at least one additional recursion processor is a Viterbi decoder, and the two recursion processors are soft-input, soft-output decoders. More preferably, there are two additional processors coupled in parallel at the inputs of the two recursion processors, respectively. All of the recursion processors, including the additional processors, perform concurrent iterative calculations on the signal. The at least one additional recursion processor calculates a quality index of the signal for each iteration and directs a controller to terminate the iterations when the measure of the quality index exceeds a predetermined level.

The quality index is a summation of generated extrinsic information multiplied by a quantity extracted from the LLR information at each iteration. The quantity can be a hard decision of the LLR value or the LLR value itself. Alternatively, the quality index is an intrinsic signal-to-noise ratio of the signal calculated at each iteration. In particular, the intrinsic signal-to-noise ratio is a function of the quality index added to a summation of the square of the generated extrinsic information at each iteration. The intrinsic signal-to-noise ratio can be calculated using the quality index with the quantity being a hard decision of the LLR value, or the intrinsic signal-to-noise ratio is calculated using the quality index with the quantity being the LLR value. In practice, the measure of the quality index is a slope of the quality index taken over consecutive iterations.

The key advantages of the present invention are easy hardware implementation and flexibility of use. In other words, the present invention can be used to stop iteration at any SISO decoder, or the iteration can be stopped at half cycles. In addition, SNR is derived according to Viterbi decoding which does not use a square root operation which would require much increased circuit

complexity or the use of an approximation. In contrast, the present invention has a very simple hardware implementation.

FIG. 6 shows simulation results using the turbo decoding in accordance with the present invention. The performance of  $Q_H(iter, \{m_i\}, L)$  and  $Q_S(iter, \{m_i\}, L)$  were verified through numerical simulations. The simulation results are presented to demonstrate the asymptotic behavior of these indexes. Then the performance of the turbo decoder is shown given that the hard and soft indexes are being used as an iteration stopping criteria. The code used is the CDMA2000 standard code with code rate 1/3,  $G_1=13$  and  $G_2=15$ , as recognized in the art. The simulation was run with 2000 frames of size 640 bits and the SNR points are 0.8dB, 0.9dB and 1.0 dB. Viterbi's memory cutting technique, known in the art for more realistic results, is implemented with synchronization learning length 30. The asymptotic behavior of the hard quality index  $Q_H(iter, \{m_i\}, L)$  and the soft quality index  $Q_S(iter, \{m_i\}, L)$  is depicted in FIGs. 6 and 7.

FIGs. 6 and 7 shows that the slope of the asymptotic curves increases as the SNR gets higher. This is as expected since high SNR gives better extrinsic information in decoding. As can be seen, the quality indexes reach their asymptotes faster as SNR increases which means less iterations are needed for convergence. This can be seen through Viterbi decoder analysis.

Based on the asymptotic behavior of the quality indexes, the decoding iterations are stopped by checking the percentage of increase (i.e. the curve slope or derivative) of these indexes. The stopping criteria used in the following plots are based on the curve slope with threshold 0.03 dB. That is if

$$\{Q_H(iter + 1, \{m_i\}, L) - Q_H(iter, \{m_i\}, L)\} / Q_H(iter, \{m_i\}, L) < 0.03 \text{ dB}$$

the iterations can be stopped. Similarly, for the soft quality index, the same threshold of 0.03 dB can be used. That is if

$$\{Q_S(iter + 1, \{m_i\}, L) - Q_S(iter, \{m_i\}, L)\} / Q_S(iter, \{m_i\}, L) < 0.03 \text{ dB}$$

the iterations are stopped. In addition, iterations can be stopped once these indexes pass a predetermined threshold to avoid any false indications.



Alternately, a certain number of mandatory iterations can be imposed before the indexes are used as criteria for iteration stopping.

The BER performance curves based on the direct use of the hard and soft quality indexes are presented in FIG. 8. In this preferred case, a mandatory  
 5 minimum number of nine iterations (that is 4.5 full cycles) are used followed by application of the quality indexes. The indexes were applied with an increment threshold 0.03 dB. The maximum number of iterations used in both cases was sixteen. The minimum number of iterations used in both cases was nine. Table 1 lists the average number of iterations to show the computation savings.

Table 1

SNR	Average number of iteration with the hard quality index	Average number of iteration with the soft quality index
0.8 dB	12.2910	13.2020
0.9 dB	11.8265	12.7995
1.0 dB	11.4195	12.3955

As expected, the number of average iterations needed decreases as SNR  
 15 increases. In addition, the degradation of signal integrity due to iteration stopping is much less than 0.1dB.

In the present invention, intrinsic SNR can also be used as an iteration stopping criteria. Due to the close relationship between intrinsic SNR and the quality indexes, the numerical results with the hard and soft intrinsic SNR has  
 20 similar performance, as shown in FIGs. 9 and 10. Only the asymptotic behavior of the sum of the last two terms in the intrinsic SNR expression are shown.

Optionally, the quality indexes and intrinsic SNR can be used as a retransmit criteria in an ARQ system. For example, using a lower threshold for frame quality, if the quality indexes or intrinsic SNR are still below the lower  
 25 threshold after a predetermined number of iterations, decoding can be stopped and a request sent for frame retransmission.

As should be recognized, the hardware needed to implement quality indexes for iteration stopping is extremely simple. Since there are LLR and extrinsic information output in each constituent decoding stage, only a MAC

(multiply and accumulate unit) is needed to calculate the soft index. Also, only one memory unit is needed to store the index and to compare it with the next one for slope calculation. Moreover, all the quality index values at different iteration stages can be stored with very few memory elements. A comparison unit based  
5 on one subtraction and one division is needed. For the hard index, a slicer is needed for hard decision before the MAC. Advantageously, these indexes can be implemented with some simple attachment to the current design.

FIG. 11 shows a flow chart representing a method 100 of terminating iteration calculations in the decoding of a received convolutionally coded signal using quality index criteria, in accordance with the present invention. A first step  
10 102 is providing a turbo decoder with two recursion processors connected in an iterative loop, and at least one additional recursion processor coupled in parallel at the inputs of at least one of the recursion processors. All of the recursion processors concurrently performing iteration calculations on the signal. In a  
15 preferred embodiment, the at least one additional recursion processor is a Viterbi decoder, and the two recursion processors are soft-input, soft-output decoders. More preferably, two additional processors are coupled in parallel at the inputs of the two recursion processors, respectively.

A next step 104 is calculating a quality index of the signal in the at least  
20 one recursion processor for each iteration. In particular, the quality index is a summation of generated extrinsic information from the recursion processors multiplied by a quantity extracted from the LLR information of the recursion processors at each iteration. The quality index can be a hard value of a soft value. For the hard value, the quantity is a hard decision of the LLR value. For  
25 the soft value, the quantity is the LLR value itself. Optionally, the quality index is an intrinsic signal-to-noise ratio (SNR) of the signal calculated at each iteration. The intrinsic SNR is a function of an initial signal-to-noise ratio added to the quality index added to a summation of the square of the generated extrinsic information at each iteration. However, only the last two terms are useful for the  
30 quality index criteria. For this case, there are also hard and soft values for the

intrinsic SNR, using the corresponding hard and soft decisions of the quality index just described.

A next step 106 is terminating the iterations when the measure of the quality index exceeds a predetermined level. Preferably, the terminating step  
5 includes the measure of the quality index being a slope of the quality index over the iterations. In practice, the predetermined level is at a knee of the quality index curve approaching its asymptote. More specifically, the predetermined level is set at 0.03 dB of SNR. A next step 108 is providing an output derived from the soft output of the turbo decoder existing after the terminating step.

While specific components and functions of the turbo decoder for convolutional codes are described above, fewer or additional functions could be employed by one skilled in the art and be within the broad scope of the present invention. The invention should be limited only by the appended claims.

## CLAIMS

What is claimed is:

1. A method (100) of terminating iteration calculations in the decoding of a received convolutionally coded signal using quality index criteria, the method comprising the steps of:
  - 5 providing (102) a turbo decoder with two recursion processors connected in an iterative loop, and at least one additional recursion processor coupled in parallel at the inputs of at least one of the recursion processors, all of the recursion processors concurrently performing iteration calculations on the signal;
  - 10 calculating (104) a quality index of the signal in the at least one recursion processor for each iteration;
  - terminating (106) the iterations when the measure of the quality index exceeds a predetermined level; and
  - providing (108) an output derived from the soft output of the turbo decoder
  - 15 existing after the terminating step.
2. The method of claim 1, wherein the first providing step includes the at least one additional recursion processor being a Viterbi decoder, and the two recursion processors are soft-input, soft-output decoders.
- 20 3. The method of claim 1, wherein the first providing step includes two additional processors being coupled in parallel at the inputs of the two recursion processors, respectively.
4. The method of claim 1, wherein the calculating step includes the quality index being a summation of generated extrinsic information multiplied by a quantity extracted from the LLR information at each iteration.

5. The method of claim 4, wherein the calculating step includes the quantity being a hard decision of the LLR value.

6. The method of claim 4, wherein the calculating step includes the quantity being the LLR value.

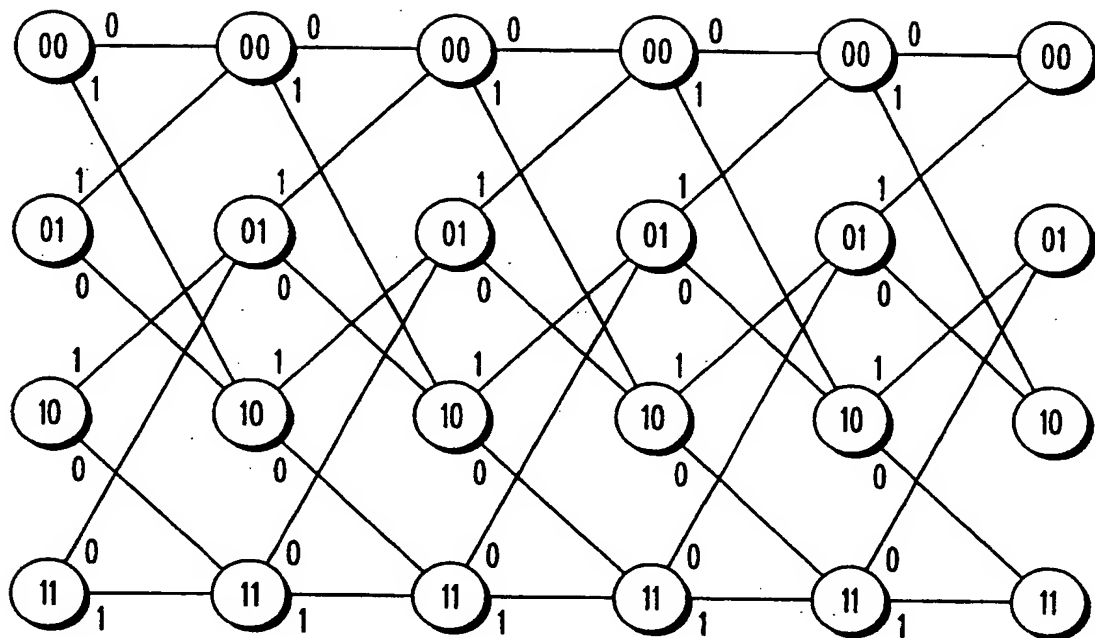
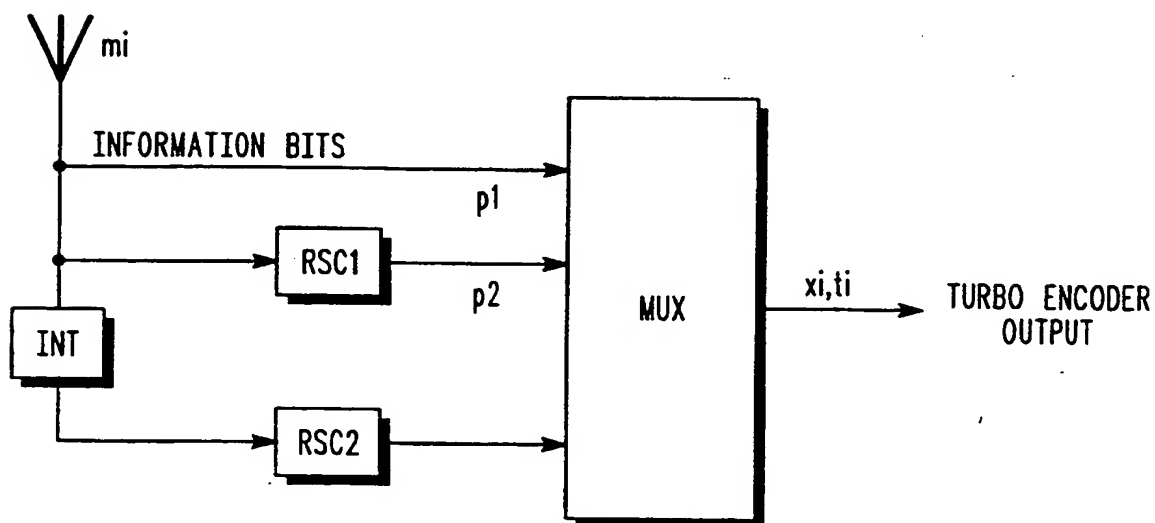
7. The method of claim 4, wherein the calculating step includes the quality index being an intrinsic signal-to-noise ratio of the signal calculated at each iteration, the intrinsic signal-to-noise ratio being a function of the quality index added to a summation of the square of the generated extrinsic information at each iteration.

8. The method of claim 7, wherein the calculating step includes the intrinsic signal-to-noise ratio being calculated using the quality index with the quantity being a hard decision of the LLR value.

9. The method of claim 7, wherein the calculating step includes the intrinsic signal-to-noise ratio being calculated using the quality index with the quantity being the LLR value.

10. The method of claim 1, wherein the terminating step includes the measure of the quality index being a slope of the quality index over the iterations.

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*FIG. 1**FIG. 2*

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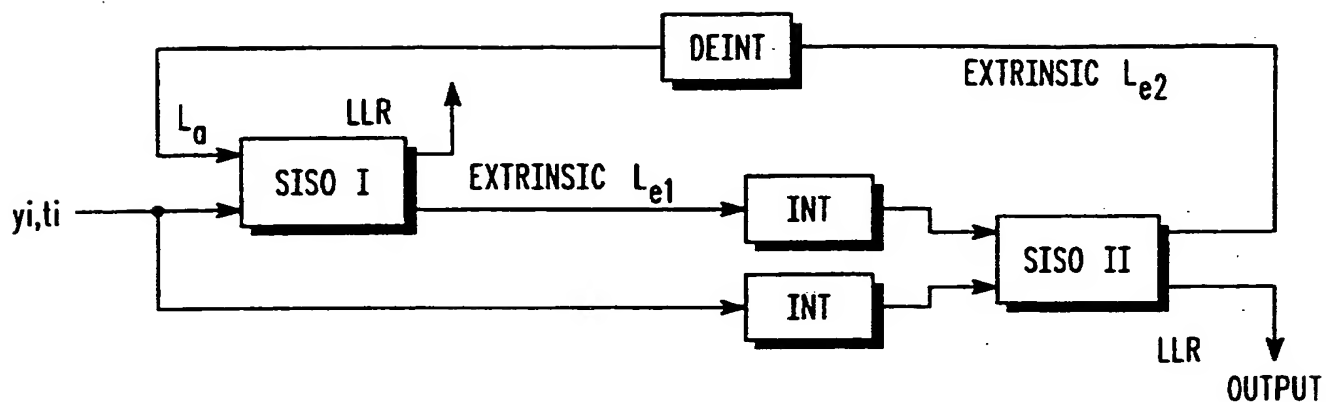


FIG. 3

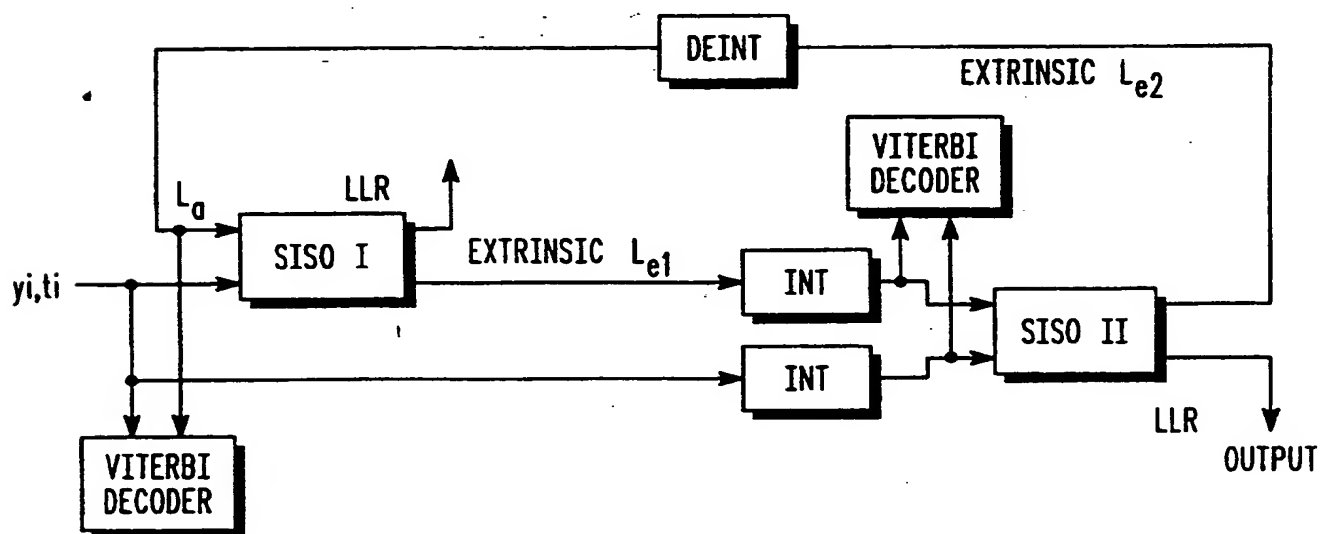


FIG. 4

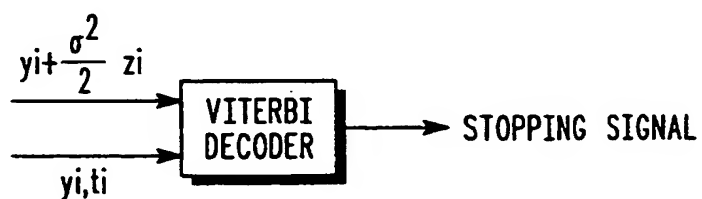
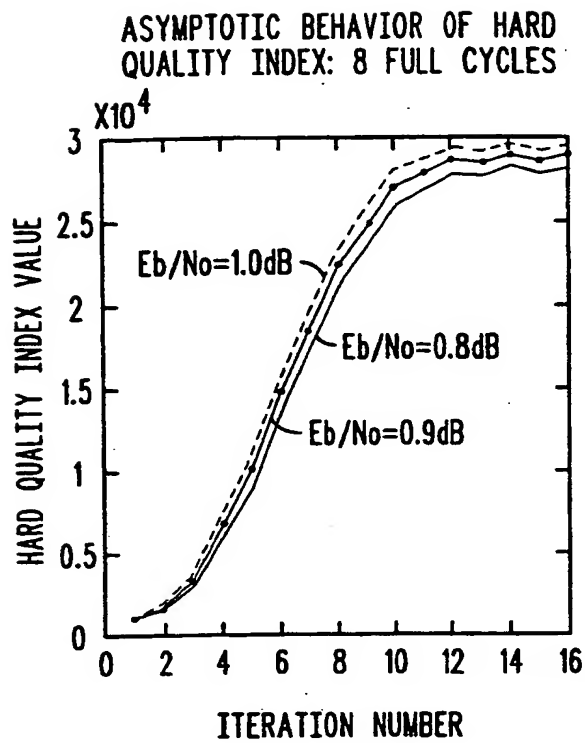
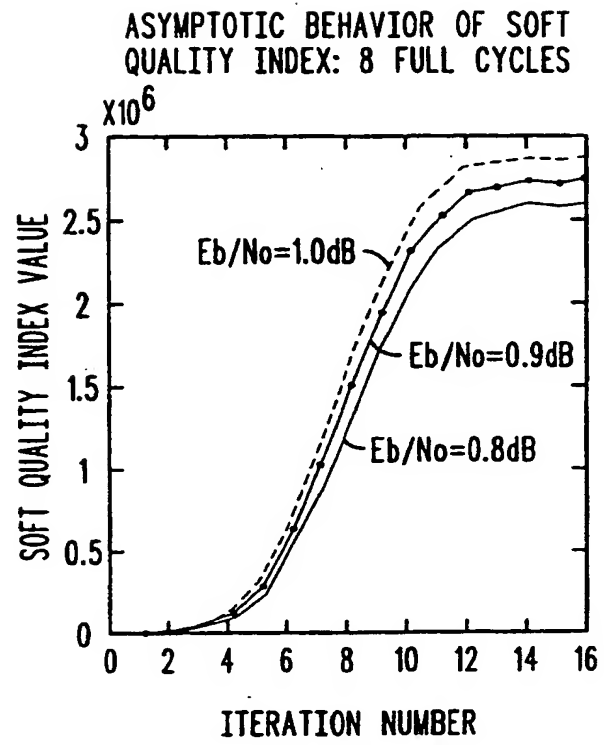
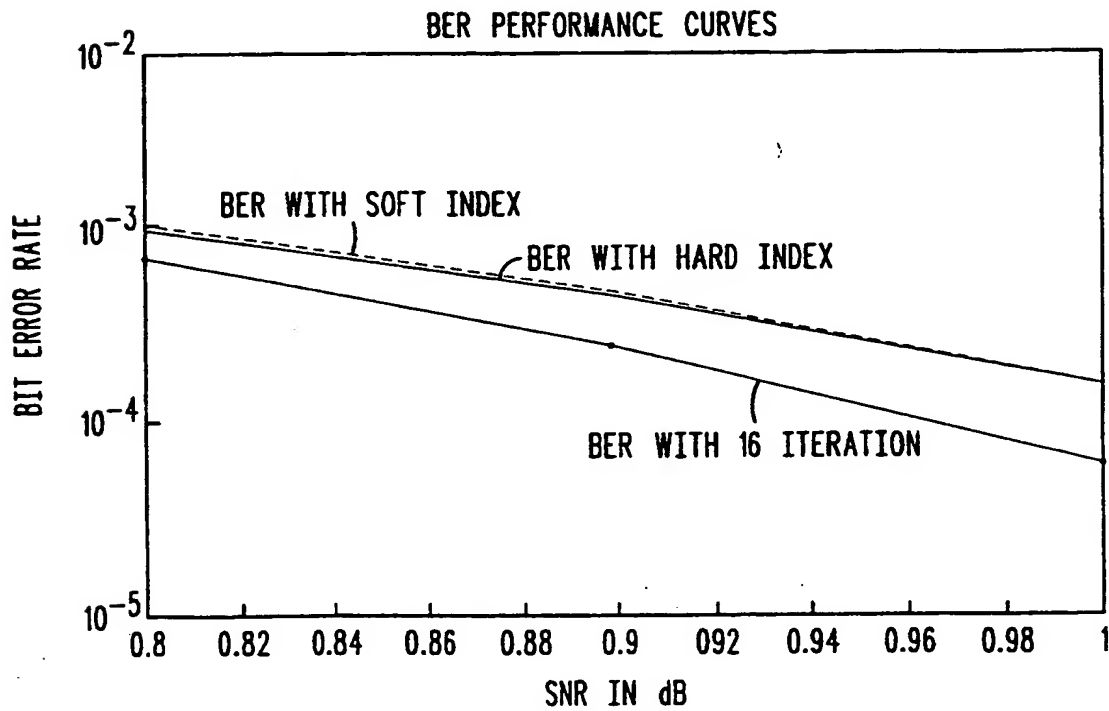


FIG. 5

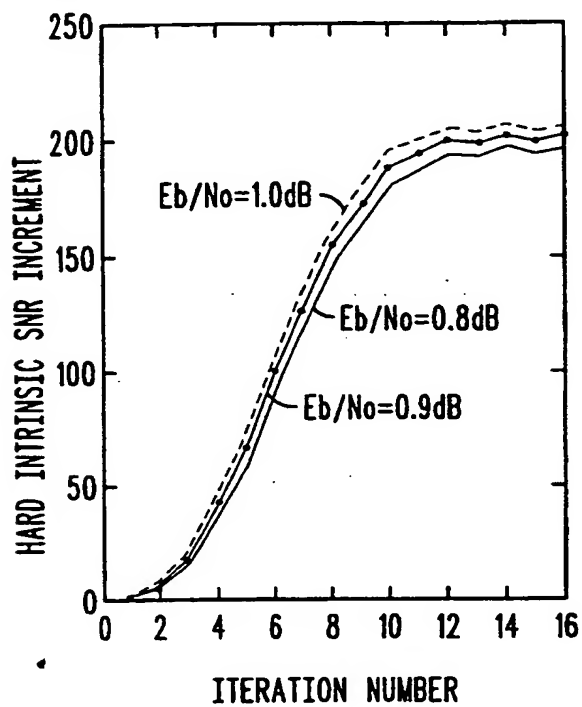
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**FIG. 6****FIG. 7****FIG. 8**

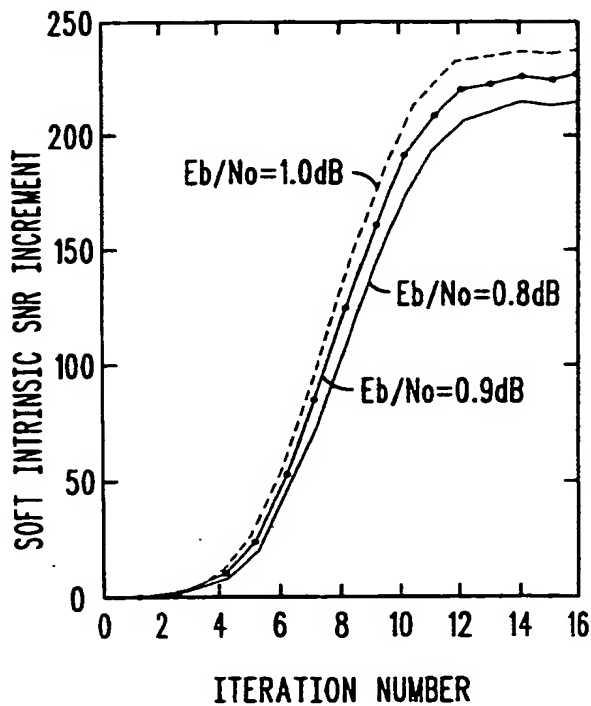


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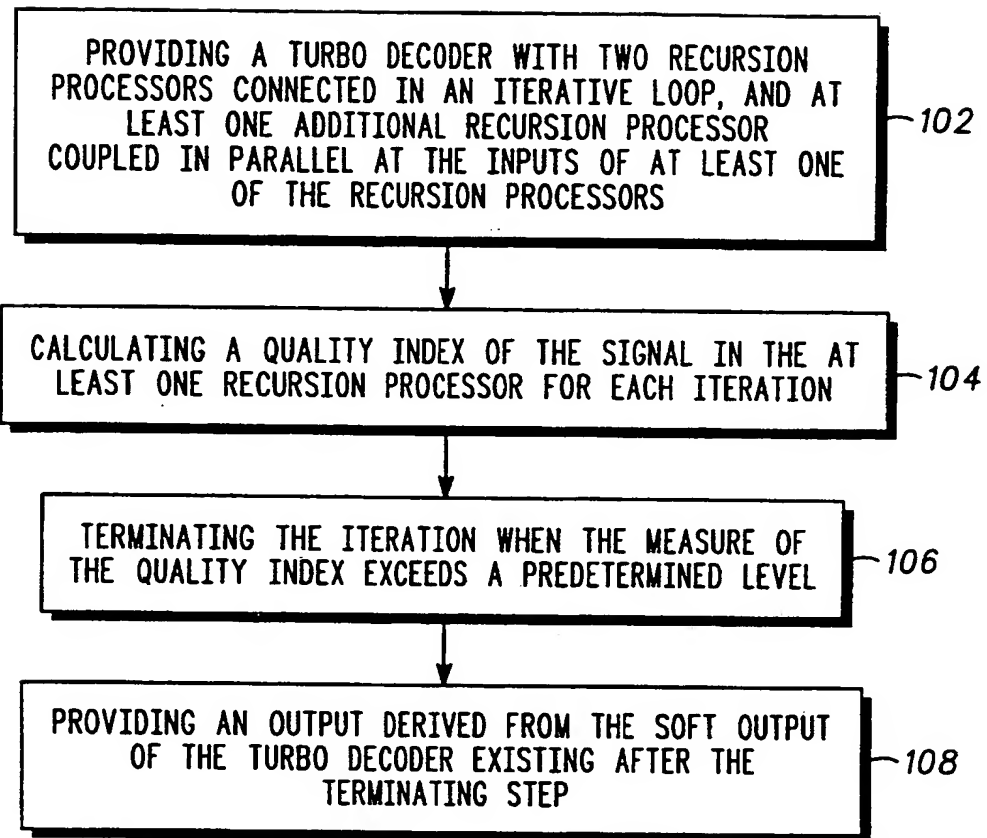
## INTRINSIC SNR INCREMENT: HARD VERSION

**FIG. 9**

## INTRINSIC SNR INCREMENT: SOFT VERSION

**FIG. 10**

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100**FIG. 11**

# INTERNATIONAL SEARCH REPORT

International application No.

PCT/US01/11544

## A. CLASSIFICATION OF SUBJECT MATTER

IPC(7) : H03M 13/00; H04L 27/06

US CL : 714/786, 794, 795; 375/340, 341

According to International Patent Classification (IPC) or to both national classification and IPC

## B. FIELDS SEARCHED

Minimum documentation searched (classification system followed by classification symbols)

U.S. : 714/786, 794, 795; 375/340, 341

Documentation searched other than minimum documentation to the extent that such documents are included in the fields searched  
NPL, IEEE, IEICE TRANS

Electronic data base consulted during the international search (name of data base and, where practicable, search terms used)  
West

## C. DOCUMENTS CONSIDERED TO BE RELEVANT

Category *	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
Y	WANG et al. Iterative (Turbo) Soft Interference Cancellation and Decoding for Coded CDMA IEEE July 1999, Vol 47. No. 7, pages 1046-1061, especially pages 1048-1052.	1-10
Y,P	CHIU, Mao-Ching Decision Feedback Soft-Input-Output Multiuser Detector for Iterative Decoding of Coded CDMA IEEE January 2001 Vol 50. No. 1, pages 25-33	4-9
Y	US 5,761,248 A (HAGENAUER et al.) 02 June 1998, col. 5-7	1-10
A	OGIWARA et al. Iterative Decoding of Serially Concatenated Punctured Trellis-Coded Modulation IEICE Trans. October 1999 Vol. E82-A No. 10, pages 2089-2095	1-10

☐ Further documents are listed in the continuation of Box C.

☐ See patent family annex.

* Special categories of cited documents:		"T"	later document published after the international filing date or priority date and not in conflict with the application but cited to understand the principle or theory underlying the invention
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"P"	document published prior to the international filing date but later than the priority date claimed		

Date of the actual completion of the international search

30 May 2001 (30.05.2001)

Date of mailing of the international search report

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Name and mailing address of the ISA/US

Commissioner of Patents and Trademarks  
Box PCT  
Washington, D.C. 20231

Facsimile No. (703)305-3230

Authorized officer

Albert Decady

Telephone No. 703-305-9600

